

IN THE CLAIMS

Please cancel claims 30-60 without prejudice.

1. (Original) A method for processing audio data, comprising:
applying a plurality of anchor models to the audio data;
mapping the output of the plurality of anchor models into frame tags; and
producing the frame tags;
wherein the plurality of anchor models comprise a discriminatively-trained classifier.
2. (Original) The method as set forth in claim 1, wherein the discriminatively-trained classifier comprises a convolutional neural network classifier.
3. (Original) The method as set forth in claim 2, further comprising training the convolutional classifier on data separate from audio data available in a use phase.
4. (Original) The method as set forth in claim 2, wherein the convolutional neural network classifier is a time-delay neural network (TDNN) classifier.
5. (Original) The method as set forth in claim 4, further comprising training the TDNN classifier using cross entropy.
6. (Original) The method as set forth in claim 1, further comprising pre-processing the audio data to generate input feature vectors for the discriminatively-trained classifier.
7. (Original) The method as set forth in claim 1, further comprising normalizing a feature vector output of the discriminatively-trained classifier.

8. (Original) The method as set forth in claim 7, wherein the normalized feature vectors are vectors of unit length.

9. (Original) The method as set forth in claim 1, further comprising:
accepting a plurality of input feature vectors corresponding to audio features contained in the audio data; and
applying the discriminatively-trained classifier to the plurality of input feature vectors to produce a plurality of anchor model outputs.

10. (Original) The method as set forth in claim 1, wherein the mapping comprises:

clustering anchor model outputs from the discriminatively-trained classifier into separate clusters using a clustering technique; and
associating a frame tag to each separate cluster.

11. (Original) The method as set forth in claim 10, further comprising applying temporal sequential smoothing to the frame tag using temporal information associated with the anchor model outputs.

12. (Original) The method as set forth in claim 1, further comprising:
training the discriminatively-trained classifier using a speaker training set containing a plurality of known speakers; and
pre-processing the speaker training set and the audio data in the same manner to provide a consistent input to the discriminatively-trained classifier.

13. (Original) A computer-readable medium having computer-executable instructions for performing the method recited in claim 1.

14. (Original) A computer-implemented process for processing audio data, comprising:
applying a plurality of anchor models to the audio data;

mapping the output of the anchor models into frame tags; and
producing the frame tags;
wherein the plurality of anchor models comprise a discriminatively-trained
classifier that is previously trained using a training technique.

15. (Original) The computer-implemented process of claim 14, wherein the
training technique employs a cross-entropy cost function.

16. (Original) The computer-implemented process of claim 14, wherein the
training technique employs a mean-square error metric.

17. (Original) The computer-implemented process of claim 14, wherein the
discriminatively-trained classifier comprises a convolutional neural network classifier.

18. (Original) The computer-implemented process of claim 14, further
comprising:
omitting an output non-linearity, which was used during training, from the
discriminatively-trained classifier to generate a modified feature vector output; and
normalizing the modified feature vector output.

19. (Original) The computer-implemented process of claim 18, wherein
normalizing further comprises creating a modified feature vector output having unit length.

20. (Original) A method for processing audio data containing a plurality of
speakers, comprising:
applying a plurality of anchor models to the audio data;
mapping an output of the anchor models into frame tags; and
constructing a list of start and stop times for each of the plurality of speakers
based on the frame tags;

wherein the plurality of anchor models comprise a discriminatively-trained classifier previously trained using a training set containing a set of training speakers, and wherein the plurality of speakers is not in the set of training speakers.

21. (Original) The method as set forth in claim 20, wherein the discriminatively trained classifier is a time-delay neural network (TDNN) classifier.

22. (Original) The method as set forth in claim 20, further comprising normalizing a feature vector output from the convolutional neural network classifier by mapping each element of the feature vector output to a unit sphere such that the feature vector output has unit length.

23. (Original) One or more computer-readable media having computer-readable instructions thereon which, when executed by one or more processors, cause the one or more processors to implement the method of claim 20.

24. (Original) A computer-readable medium having computer-executable instructions for processing audio data, comprising:

training a discriminatively-trained classifier in a discriminative manner during a training phase to generate parameters that can be used at a later time by the discriminatively-trained classifier;

applying the discriminatively-trained classifier that uses the parameters to the audio data to generate anchor model outputs; and

clustering the anchor model outputs into frame tags of speakers that are contained in the audio data.

25. (Original) The computer-readable medium of claim 24, further comprising pre-processing a speaker training set during the training and validation phase to produce a first set of input feature vectors for the discriminatively-trained classifier.

26. (Original) The computer-readable medium of claim 25, further comprising pre-processing the audio data during the use phase to produce a second set of input feature vectors for the discriminatively-trained classifier, the pre-processing of the audio data being preformed in the same manner as the pre-processing of the speaker training set.

27. (Original) The computer-readable medium of claim 24, further comprising normalizing the feature vector outputs to produce feature vectors having a unit length.

28. (Original) The computer-readable medium of claim 27, wherein normalizing further comprises omitting a nonlinearity from the discriminatively-trained classifier during the use phase.

29. (Original) The computer-readable medium of claim 25, further comprising applying temporal sequential smoothing to the clustering the clustered feature vector outputs to produce the frame tags.

Claims 30-60: Canceled